

DOC FILE COPY.

ARL-TM-79-3



(10) & Copy No. 47

EXPERIMENTAL SUPERPOSITION OF ACOUSTIC FIELDS FOR DETECTION PERFORMANCE STUDIES

Jack A. Shooter Milton L. Gentry, Jr.

APPLIED RESEARCH LABORATORIES
THE UNIVERSITY OF TEXAS AT AUSTIN
POST OFFICE BOX 8029, AUSTIN, TEXAS 78712

15 March 1979

Technical Memorandum

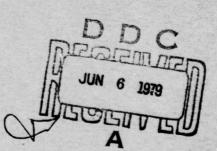
APPROVED FOR PUBLIC RELEASE; DISTRIBUTION UNLIMITED.

Prepared for:

NAVAL ELECTRONIC SYSTEMS COMMAND DEPARTMENT OF THE NAVY WASHINGTON, DC 20360







SECURITY CLASSIFICATION OF THIS PAGE (When Date Entered)

REPORT DOCUMENTAT		READ INSTRUCTIONS BEFORE COMPLETING FORM
REPORT NUMBER	2. GOVT ACCESSION NO.	3. RECIPIENT'S CATALOG NUMBER
TITLE (and Subtitle)		5. TYPE OF REPORT & PERIOD COVERED
EXPERIMENTAL SUPERPOSITION OF		technical memorandum
FOR DETECTION PERFORMANCE STU	DIES .	ARL-TM-79-3
AUTHOR(a)		. CONTRACT OR GRANT NUMBER(4)
Jack A./Shooter Milton L./Gentry, Jr/	(15	N00039-78-C-0307
PERFORMING ORGANIZATION NAME AND AD Applied Research Laboratories	DRESS	10. PROGRAM ELEMENT, PROJECT, TASK
The University of Texas at Au Austin, Texas 78712		(12) 34p.
. CONTROLLING OFFICE NAME AND ADDRES Naval Electronic Systems Comm	/	15 March 1979
Department of the Navy Washington, DC 20360		33
MONITORING AGENCY NAME & ADDRESS(II	different from Controlling Office)	15. SECURITY CLASS. (of this report)
		UNCLASSIFIED
		15. DECLASSIFICATION/DOWNGRADING SCHEDULE N/A
7. DISTRIBUTION STATEMENT (of the abatract	entered in Block 20 II different fro	m Report)
7. DISTRIBUTION STATEMENT (of the abatract	entered in Block 20, it dilierent tro	т керогі)
B. SUPPLEMENTARY NOTES		
. KEY WORDS (Continue on reverse side if neces	neary and identify by block number)	
•		
superposition		
superposition		- 434 Du
\	404	434 Du
U) The concept of superimposin noise acoustic field is dis	g a target acoustic focussed and sample reserved. The purpose	field onto an independent sults are presented using is to explore the
U) The concept of superimposin noise acoustic field is dis data from the deep Atlantic principles and methods of h	g a target acoustic for the cussed and sample reserved to the concept can be sence of noise as appreciate the customers.	field onto an independent sults are presented using is to explore the se implemented for assessing clied to particular systems.

SECURITY CLASSIFICATION OF THIS PAGE(When Date Entered)

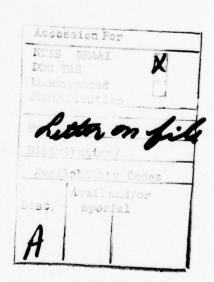
20. Cont'd.

scaled down in amplitude and reinserted into the noise spectrum. The detection results of the actual low level signal and the reconstructed low level signal are compared and the results are shown to be essentially the same.

UNCLASSIFIED

TABLE OF CONTENTS

		Page
I.	INTRODUCTION	1
II.	GENERAL PROBLEM	3
III.	SUPERPOSITION	5
IV.	IV. EXPERIMENTAL TEST ON OMNI DATA	
	A. A Met d for Adjusting a High Level Signal	12
	B. The Experiment	12
	C. Data Processing	13
	D. Results	16
REFE	RENCES	27



LIST OF FIGURES

Figure	<u>Title</u>	Page
1	Difference in Level Between the 156 Hz cw Signal and the 160 Hz cw Signal on Hydrophone 3 Over the First Run	15
2	3-D Spectra Plots with and without the Reconstructed Signal for the Low Frequency Sources	17
3	3-D Spectra Plots with and without the Reconstructed Signal for the High Frequency Sources	18
4	Signal Excess and Noixe versus Range for the High Frequencies Over Run ${\bf 1}$	20
5	Signal Excess and Noise versus Range for the High Frequencies Over Run 2	21
6	Signal Excess and Noise versus Range for the Low Frequencies Over Run 2	22
	LIST OF TABLES	
<u>Table</u>	<u>Title</u>	Page
I	Percent of Signal Detections Using Hydrophone 3	23
II	Percent of Signal Detections Using Hydrophone 6	24

I. INTRODUCTION

This memorandum will discuss the concept of superimposing a target acoustic field onto an independent noise acoustic field and will present some example results using data from the deep Atlantic Ocean. The purpose is to explore the principles and methods of how this concept can be implemented for assessing signal detection in the presence of noise as applied to particular systems. Section II will outline the general problem and provide the motivation for the superpositional approach to the solution. Section III will discuss the principle of superposition for a point receiver and an array of point receivers. Section IV will present experimental results where high and low level signals were used simultaneously. From these data, the high level signal was scaled down in amplitude and reinserted into the noise spectrum. The detection results of the actual low level signal and the reconstructed low level signal are compared and the results are shown to be essentially the same.

II. GENERAL PROBLEM

An acceptable procedure for assessing how well a system can detect a target signal in a noise field is to test it in a real environment over an extended period under many signal and noise conditions. This can be very expensive over an extended period of time and the environmental conditions will not always meet the specified noise conditions. The most serious problem is the general nonstationary aspect of the noise field which creates an interpretation problem in the case of a range changing source. That is, level changes in the noise field will change the signal-to-noise ratio (S/N) at the same time the signal level changes due to changing signal-receiver geometry. This means that S/N versus range and detection versus range must be very carefully assessed and interpreted in reference to the noise field. A second serious problem can be caused by the noise from a ship towing a target signal source, masking the ambient background noise. This means that in some cases, especially at short range, S/N measurements will not be signal to ambient background noise, but instead will be signal to tow ship noise.

It is proposed that a way of circumventing these problems is to record the signal and noise fields independently under field conditions. Because of the importance of boundary conditions and changes in a time varying environment, it cannot be overemphasized that the signal and noise fields must be recorded at the same site and position. It is further emphasized that the signal field repeatability should be tested to establish confidence in the stability of the final results. A second emphasis is

placed on recording the noise field over a time sufficient to measure different noise conditions. For example, it has been shown that typical noise decorrelation times in the Atlantic Ocean and Mediterranean Sea are near 10 h in the shipping frequency band of the spectrum and a maximum of 25 h in the wind dominated part of the spectrum. 1 This means that, on the average, it is desirable to have 10×25 h or 10 days worth of noise recording to cover different levels of the noise field.

III. SUPERPOSITION

This section will briefly discuss the principle of superposition and describe a procedure that may be applied to superimpose real data recorded from an omnidirectional sensor and an array of sensors.

The principle of superposition is well known and discussed in every textbook on acoustics. Most solutions to the wave equation assume linear propagation and separate the solution into its components (e.g., modes) and build a complete solution by the principle of superposition. In effect most signal processing analyses assume linear systems and separate the acoustic field into its components (e.g., multiple source signals plus multiple noise sources) for ease of manipulation.

The basic assumption is that the acoustic pressure field at a point in space is the sum of pressure generated from multiple independent sources. Therefore they may be treated independently. In this case we want to record a high level signal field and sum it into one or more noise fields at different relative levels. Once again it is emphasized that this must be done with in situ real data because of the importance of boundaries and the many paths that a signal can take to arrive at a point. That is, computer simulated data would not be acceptable. In principle the recorded data to be scaled and summed can be in analog or digital form. Because of the experience and background of Applied Research Laboratories, The University of Texas at Austin (ARL:UT), in digital signal processing, the digital processing techniques would be the easiest and most accurate to implement.

The most straightforward example is in the case of a single omni sensor. In the case of independent signal and noise fields, the data can be processed independently before detection. That is, the signal spectra and noise spectra can be computed independently and stored on magnetic tape, just as the standard ARL:UT AN/cw processor currently operates on data. The superposition then takes place in the frequency domain by scaling the signal spectrum to match a desired source level and summing onto a noise spectrum chosen to represent a particular noise condition. This is described in the following set of simple equations.

Let s be the received signal from a known source of known level and position and let n be an independent noise sample received by the same sensor in the same position but at a different time. The summed field is shown in Eq. (1) where the signal field is scaled by a.

$$x(t) = as(t) + n(t)$$
 (1)

It is possible to make this sum directly in the time domain given the sampled time series s and n. However, the most costly and time consuming part of the ARL:UT processing is in the FFT time. If the sum is to be repeated for several values of a corresponding to different source levels and also repeated for different noise fields, then it would be far more efficient if the FFT's had to be done only once. It can be shown that it is necessary to do the FFT's only once for a specified set of parameters giving frequency resolution and averaging times (ALI's) by Fourier transforming Eq. (1) and forming the ensemble average of the resulting power spectra. Let $X(\omega)$ be the complex Fourier transform of x(t) in Eq. (1), $S(\omega)$ be the transform of s(t), and $N(\omega)$ be the transform of n(t). Then the Fourier transform of Eq. (1) becomes

$$X(\omega) = aS(\omega) + N(\omega), \qquad (2)$$

and the energy spectrum of Eq. (2) becomes

$$X(\omega)X^*(\omega) = a^2S(\omega) S^*(\omega) + N(\omega)N^*(\omega) + S(\omega)N^*(\omega) + N(\omega)S^*(\omega).$$
 (3)

The ensemble average of Eq. (3) is

$$\langle XX*\rangle = a^2 \langle SS*\rangle + \langle NN*\rangle, \tag{4}$$

where

$$\langle SN* \rangle = \langle NS* \rangle = 0$$
,

because S and N are independent and zero biased in time and frequency.

Therefore Eq. (4) means that the signal and noise fields can be processed separately into spectra and stored, e.g., on tape, for postprocessing, using whatever scale factor a² is desired. For example, the postprocessing might involve frequency detection and line tracking as part of the signal classification.

The next case to be considered is that of an array of sensors to accomplish spatial processing. In this case each sensor can be treated independently in the time domain by sampling the signal and noise fields at different times, storing the time series, scaling the signal field, summing, and then forming beams in the frequency domain. Once again this technique is inefficient if a parameter study is to be done because of the number of times the beamforming and FFT process would have to be redone. Fortunately the same principle used in the single omnidirectional case can be applied to an array. This means that the signal field and noise fields can be processed independently and stored in the form of complex cross-spectral matrices, and the matrices can be later summed

using different scaling factors a. This can easily be shown to be valid by considering that, in spatial processing, the spatial information is contained in the differences between elements. Therefore it is necessary to consider only the i and j elements in an array, as shown in Eq. (5).

$$x_{i}(t) = s_{i}(t) + n_{i}(t)$$
. (5)
 $x_{j}(t) = s_{j}(t) + n_{j}(t)$.

This represents the superposition of the S and N fields at spatial points "i" and "j". The Fourier transform of Eq. (5) is

$$X_{i}(\omega) = S_{i}(\omega) + N_{i}(\omega)$$

$$X_{j}(\omega) = S_{j}(\omega) + N_{j}(\omega),$$
(6)

and the cross-spectra between i and j is seen in Eq. (7).

$$X_{i}X_{j}^{*} = S_{i}S_{j}^{*} + S_{i}N_{j}^{*} + N_{i}S_{j}^{*} + N_{i}N_{j}^{*}$$
 (7)

The ensemble average of Eq. (7) is

$$\langle x_i x_j^* \rangle = \langle s_i s_j^* \rangle + \langle v_i v_j^* \rangle,$$
 (8)

where

$$\langle S_i N_i^* \rangle = \langle N_i S_i^* \rangle = 0.$$

This means that, for independent signal and noise fields, the data can be processed independently into cross-spectral matrices which contain all information necessary for beamforming. Scaling and superposition can be applied to the matrices directly. In the case of a line array, the matrices

can be further compressed for data storage by averaging the diagonal elements (complex) and storing only those terms, which may be called the spatial correlation samples. Given the summed or superimposed matrices the post-processing can then form beams and do signal detection and tracking as in a normal processor.

IV. EXPERIMENTAL TEST ON OMNI DATA

A test of the theory outlined in section III was implemented using archived data from the deep Atlantic Ocean. In this exercise a cw source that generated high and low level signals simultaneously was used. The generated signals were the frequency pairs 70 Hz, 68 Hz and 156 Hz, 160 Hz and they were collocated. Collocation is especially important in data interpretation and analysis. It is well known that slight variations in source receiver geometry (e.g., source depth variations) will cause large fluctuations in signal level. The fact that they were collocated removes the effects of geometry between the high and low level signals.

The data were recorded by the ommidirectional hydrophones on an ACODAC and reduced to 5 min average spectra at a frequency resolution of 0.073 Hz. The archived spectra were then used to experimentally test the detection differences between an actual low level signal and a reconstructed low level signal. In this case the high level signal was scaled down and shifted in frequency before it was summed back into the noise field. In a real simulation exercise, where high level signals are to be used to reconstruct low level signals in a different noise field, the frequency shift would not be necessary. The advantage of these data is that the noise field is essentially the same for the real and synthesized signals, and the sources were collocated. Except for slight variations in noise because of different frequency bands, the differences in detection performance can be compared directly.

The experimental results here are limited to omnidirectional data.

The theory in section III accounts for array processing but a definitive experimental test will have to be conducted at a later time.

The following sections discuss the details of the procedures used to obtain signal excess as a function of range for a real low level signal and a reconstructed low level signal.

A. A Method for Adjusting a High Level Signal

Given a power spectrum, i.e., a set of Fourier coefficients modulo squared, containing a high level cw signal, the first step in the procedure is to rescale those coefficients to correspond to the desired low level cw signal. The coefficients can then be added back into the spectrum in another band.

For example, suppose $\{C_{\nu}^{i+j}\}_{\nu=1}^{i+j}$ are the ith through i+jth FFT coefficients of a band containing a signal. The signal contained within this band can be scaled by "a" and moved to another band, $\{C_{\nu}^{k+j}\}_{\nu=k}^{k+j}$ by this simple process:

$$C_{\nu} = C_{\nu} + a^2 C_{i+\nu-k}$$
 for $\nu = k, k+1, ..., k+j$. (9)

The effect is to change the level of the signal by

$$\Delta SL = 10 \log a^2 \tag{10}$$

and to shift the signal to another frequency. In the data chosen, the noise carried along in the signal band will be scaled down by the same amount a^2 , and when it is summed back into the spectrum it will have negligible effects.

B. The Experiment

The experiment performed was to take high level cw signals that were transmitted during the exercise and scale them to

transmitted. In particular, 70 Hz and 160 Hz high level signals were transmitted simultaneously with low level 68 Hz and 156 Hz signals.

The high level 70 Hz and 160 Hz signals were scaled and shifted to 66 Hz and 166 Hz, respectively. Comparisons of signal excess were made between the reconstructed 66 Hz and real 68 Hz signals as well as between the real 156 Hz and the reconstructed 166 Hz signals.

The spectra used for analysis were taken from two ACODAC hydrophones, No. 3 and No. 6. Hydrophone 3 was at a depth of 4750 m, which corresponded to critical depth. Hydrophone 6 was at the bottom, or a depth of 5250 m.

The spectra were 5 min averages with a frequency range from 0 to 300 Hz and a coefficient spacing of 0.073 Hz.

Data from two radial cw tows were utilized for the high frequency signals. Run 1 was a southwest run away from the ACODAC and Run 2 was a north-northeast run also away from the receiver. The low frequency data were taken from the second run only.

C. Data Processing

The actual data processing consisted of determining the adjustments to the high level signals; adjusting the high level signals; computing background noise level estimates, signal excesses, and signal levels for the various cw signals; and finally, displaying the results.

Rather than rely on absolute source levels it was decided that the significant parameters were the relative differences in source level between the high and low level signals. The adjustment to the high level signal was estimated as the median difference in level between the high and low level source pairs. This was measured when the sources were close to the receiver (<20 km). An example of measured differences in received SPL is shown in Fig. 1.

Figure 1 is shown on a scale of 5 dB per increment. A scatter of 2 dB is due to variance in the noise estimates used to debias the signal level estimates. The additional variations are due to received level differences between the two frequencies. The adjustment for the 160 Hz source was found to be -15.0 dB for Run 1 and -14.5 dB for Run 2. The median level difference between the 70 Hz and 60 Hz sources was found to be -22.0 dB (only the second tow was used). The median difference in background noise levels between the 68 Hz and the 66 Hz signals was found to be 0.55 dB; thus the final adjustment to the 70 Hz signal was chosen to be -21.45 dB. This was done so that the signal excess plots of the 68 Hz and 66 Hz signals could be compared. The noise level differences between 156 Hz and 166 Hz were considerably smaller and were ignored.

With the adjustment determined, the high level 70 Hz and 160 Hz signals were reduced in level, shifted in frequency to 66 Hz and 166 Hz, respectively, and then superimposed back onto the spectrum.

The noise level N around a given cw signal was a percentile estimate based on the FFT coefficients contained in a 3.0 to 3.5 Hz wide band centered about the cw signal, deleting the coefficients contained in a 1 Hz band centered about the cw signal.

Coefficient sums around the peak of 1, 3, and 5 elements were found over the wide band as well as over a smaller 0.5 Hz wide band centered about the cw signal. When the center frequencies of the wide

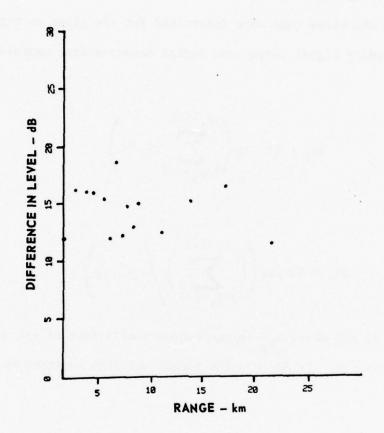


FIGURE 1

DIFFERENCE IN LEVEL BETWEEN THE 156 Hz cw SIGNAL AND
THE 160 Hz cw SIGNAL ON HYDROPHONE 3 OVER THE FIRST RUN

band peak sums were not the same as the center frequencies of the narrowband peak sums, the signal was rejected as a false detection.

Once the three sums were determined for the given cw signal, three corresponding signal levels and signal excesses were computed as

$$SL_{i} = 10 \log \left(\sum_{j=p_{i}-i-1}^{p_{i}+i-1} (C_{j}-N) \right)$$
 (11)

and

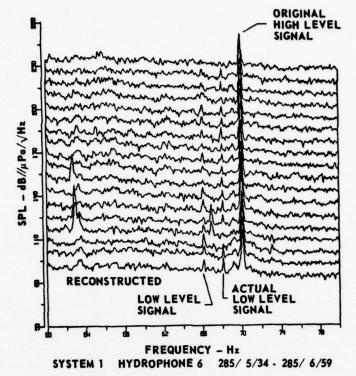
$$SE_{i} = 10 \log \left(\left(\sum_{j=p_{i}-i-1}^{p_{i}+i-1} C_{j} \right) / (2i-1)N \right)$$
 (12)

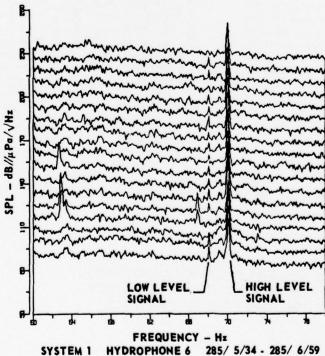
for i = 1, 2, 3, and where C_{p_i} is the center coefficient of the (2i-1) coefficient peak sum. If $SE_i < 0$, the signal was also rejected and SL_i was set to 0.

D. Results

The basic result was that the signal excess and the noise for the reconstructed and actual signals behave similarly when the plots are compared.

Figures 2 and 3 show three-dimensional spectra illustrating the insertion of a rescaled signal into a noise spectrum for the low frequency 66 Hz case and for the high frequency 166 Hz case. Both of these examples are for hydrophone 6 which was on the bottom. The signal excess is slightly higher on the bottom hydrophone because the noise for these data is slightly lower on the bottom hydrophone.





ARL:UT AS-79-428 JAS-GA 3-20-79

FIGURE 2
3-D SPECTRA PLOTS WITH AND WITHOUT
THE RECONSTRUCTED SIGNAL FOR
THE LOW FREQUENCY SOURCES

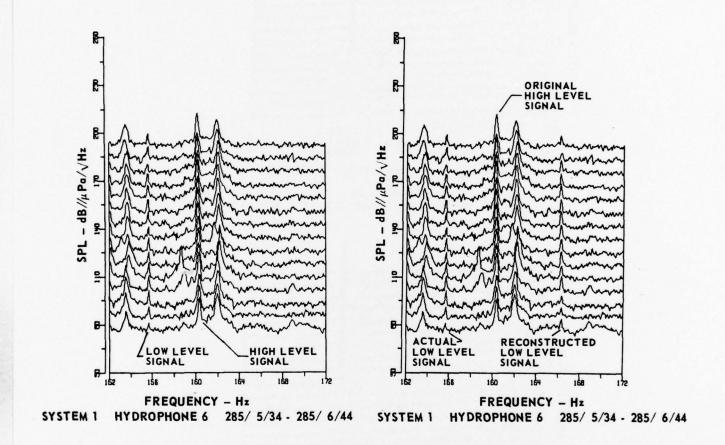


FIGURE 3
3-D SPECTRA PLOTS WITH AND WITHOUT
THE RECONSTRUCTED SIGNAL FOR
THE HIGH FREQUENCY SOURCES

ARL:UT AS-79-429 JAS-GA 3-20-79 The salient points are in the figures showing the signal excess. Figure 4 shows signal excess for the critical depth receiver (HYD 3) and the bottom receiver (HYD 6) for the real (156 Hz) and reconstructed (166 Hz) signals. The signal excess as a function of range is on the average the same in all cases.

The details differ because of slight variations in the noise field. Please note that the noise scale in the figures is compressed 2:1 relative to the signal excess. Figure 5 shows signal excess for the same frequencies at a different time and it is noted that the noise is lower on the deeper receiver. Again the signal excess is the same, on the average, as a function of range for the real and reconstructed signals. The low frequency results are shown in Fig. 6; because of the higher noise level in this part of the spectrum, the signal excess is lower. However, the comparison of signal excess between the real and reconstructed signals is again the same.

The percentage of signals detected using the method discussed in section IV.C. tended to be higher for the reconstructed signal. The reconstructed signal detection percentages ranged from 35% to 88%, whereas the actual signal detection percentage ranged from 30% to 75%. The percent detections are summarized in Tables I and II. In this test the detection algorithm was a conservative estimator because it was designed to detect and measure cw propagation loss. A different algorithm could raise the probability of detection.

As a sensitivity study, the 166 Hz signal was simulated with adjustments of -14.5 dB, -15.0 dB, and -15.5 dB. The effect of the 0.5 dB perturbations in the 166 Hz level adjustment was simply to shift the

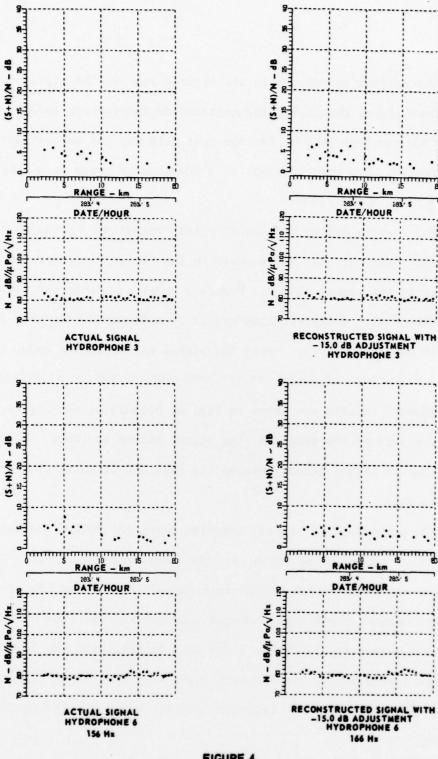


FIGURE 4
SIGNAL EXCESS AND NOISE versus RANGE
FOR THE HIGH FREQUENCIES OVER RUN 1

ARL:UT AS-79-430 JAS-GA 3-20-79

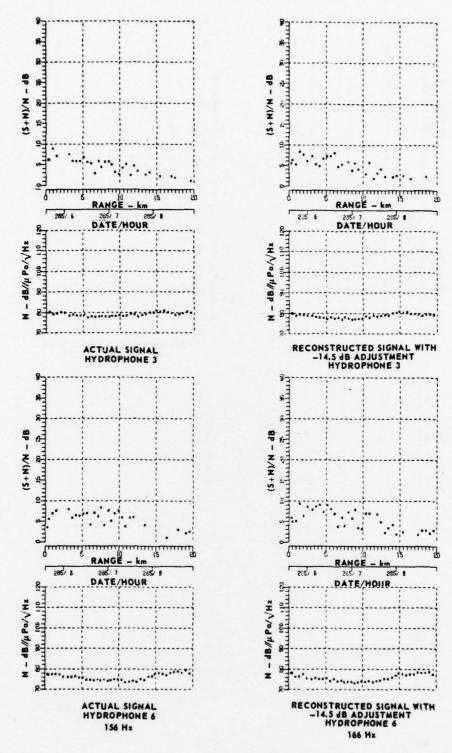


FIGURE 5
SIGNAL EXCESS AND NOISE versus RANGE
FOR THE HIGH FREQUENCIES OVER RUN 2

ARL:UT AS-79-43 JAS-GA 3-20-79

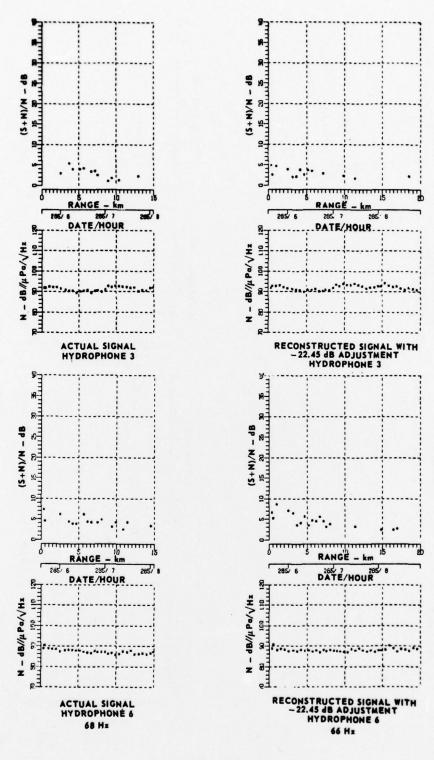


FIGURE 6
SIGNAL EXCESS AND NOISE versus RANGE
FOR THE LOW FREQUENCIES OVER RUN 2

ARL:UT AS-79-432 JAS-GA 3-20-79

TABLE I

PERCENT OF SIGNAL DETECTIONS USING HYDROPHONE 3

	REAL 156 Hz	RECONSTRUCTED 166 Hz	REAL 68 Hz	RECONSTRUCTED 66 Hz
Run 1	45	74	-	
Run 2	75	75	30	35

TABLE II

PERCENT OF SIGNAL DETECTIONS USING HYDROPHONE 6

	REAL 156 Hz	RECONSTRUCTED 166 Hz	REAL 68 Hz	RECONSTRUCTED 66 Hz
Run 1	58	74		
Run 2	70	88	40	48

signal excess curves by about the 0.5 dB increment. The small shifts in this case were not enough to change the detection statistics. This does show that fine tuning adjustments can be made in the signal excess and that proper scaling is necessary.

REFERENCES

- A. J. Perrone and L. A. King, "Analysis technique for classifying wind and ship-generated noise characteristics," J. Acoust. Soc. Am. <u>58</u>, 1186-1189 (1975).
- H. P. Bucker, "Cross sensor beamforming with a sparse line array,"
 J. Acoust. Soc. Am. 61, 494-498 (1977)

DISTRIBUTION LIST FOR ARL-TM-79-3 UNDER CONTRACT N00039-78-C-0307 UNCLASSIFIED

Copy No.

Copy No.	
	Commander
	Naval Electronic Systems Command
	Department of the Navy
	Washington, DC 20360
1	Attn: PME-124
2- 4	PME-124/30
5	PME-124/40
5 6	PME-124/60
7	PME-124TA
	Assistant Director of the Navy (RE&S)
	Washington, DC 20301
8	Attn: G.A. Cann
	Chief of Naval Operations
	Department of the Navy
	Washington, DC 20350
9	Attn: OP-095
10	OP-951
11	OP-955
	Chief of Naval Development
	Headquarters, Naval Material Command
	Washington, DC 20360
12	Attn: MAT-035
	Commander
	Naval Air Development Center
	Department of the Navy
	Warminster, PA 18974
13	Attn: Code 205
14	Code 2052
15	Code 2055
16	Code 20P1
17	Code 2604
18	Code 602
19	Chief of Naval Research
	Department of the Navy
	Arlington, VA 22217
20	Commanding Officer
	Naval Research Laboratory
	Washington, DC 20375
21	Attn: Code 8100
22	Code 8160

Copy No. 23 Oceanographer of the Navy Hoffman Building II 200 Stovall Street Alexandria, VA 22332 Commander Naval Air Systems Command Department of the Navy Washington, DC 20360 24 Attn: PMA-264 25 Commander Naval Sea Systems Command Department of the Navy Washington, DC 20362 26 Attn: Code-06H1 27 Commander Naval Ocean Systems Center San Diego, CA 92152 28 Attn: M.R. Akers 29 Dr. R.A. Wagstaff 30 Dr. R.R. Gardner 31 Commander Naval Surface Weapons Center White Oak Laboratory 8121 Georgia Avenue Silver Spring, MD 20910 32 Commander New London Laboratory Naval Underwater Systems Center New London, CT 06320 Commander Naval Oceanographic Office Department of the Navy NSTL Station, MS 39529 33 Attn: W. Geddes Commanding Officer Naval Intelligence Support Center 4301 Suitland Road

Washington, DC 20390

Attn: Code 222

34

Market Bridge Br

Copy No.	
35	Commanding Officer Naval Ocean Research and Development Activity NSTL Station, MS 39529
36	Attn: Code 320
37	Code 340
	Code 500
38	Code 600
39	Defense Advanced Research Projects Agency 1400 Wilson Boulevard Arlington, VA 22209
40	Attn: Dr. T. Kooij
40	ARPA Research Center Unit 1, Bldg. 301A NAS Moffett Field, CA 94035
41	Attn: E.L. Smith
42	Superintendent Naval Postgraduate School Monterey, CA 93940
	Antisubmarine Warfare Systems Project Office Department of the Navy Washington, DC 20360
43	Attn: PM-4
44-53	Commanding Officer and Director Defense Documentation Center Defense Services Administration Cameron Station, Building 5 5010 Duke Street Alexandria, VA 22314
54	Applied Physics Laboratory The Johns Hopkins University The Johns Hopkins Road Laurel, MD 20910 Attn: Dr. G.L. Smith
34	Acou. Di Cibi omeon
55	Applied Physics Laboratory The University of Washington 1013 NE Fortieth Street Seattle, WA 98195

Copy No. Arthur D. Little, Inc. Acorn Park Cambridge, MA 02140 Attn: Dr. G.R. Raisbeck 56 Bolt, Beranek, and Newman, Inc. 57 1701 N. Fort Myer Drive Arlington, VA 22209 Bell Telephone Laboratories 58 2 Whippany Road Whippany, NJ 07981 Sanders Associates, Inc. 24 Simon Street Nashua, NH 03060 59 Attn: L. Gagne Scripps Institution of Oceanography at The University of California, San Diego La Jolla, CA 92093 Attn: V.C. Anderson 60 Planning Systems, Inc. 7900 Westpark Drive McLean, VA 22101 61 Attn: Dr. R.L. Spooner Science Applications, Inc. 8400 Westpark Drive McLean, VA 22101 62 Attn: Dr. J.S. Hanna Sutron Corporation Suite 700 1925 N. Lynn Street Arlington, VA 22209 Attn: C.H. Dabney 63 Tetra-Tech, Inc.

1911 N. Ft. Meyer Drive Arlington, VA 22209 Attn: W.E. Sims

64

Copy No.

78-82

Tracor, Inc. 1601 Research Blvd. Rockville, MD 20850 Attn: J.T. Gottwald 65 TRW, Inc. TRW Defense & Space Systems Group Washington Operation 7600 Colshire Drive McLean, VA 22101 Attn: J. Bender 66 Underwater Systems, Inc. World Building 8121 Georgia Avenue Silver Spring, MD 20910 Attn: Dr. M.S. Weinstein 67 Western Electric Company 68 P.O. Box 20046 Greensboro, NC 27420 Woods Hole Oceanographic Institution Woods Hole, MA 02543 Attn: E.E. Hays 69 Glen E. Ellis, ARL:UT 70 Loyd D. Hampton, ARL:UT 71 Kenneth E. Hawker, ARL:UT 72 73 Stephen K. Mitchell, ARL:UT Clark S. Penrod, ARL:UT 74 75 Jack A. Shooter, ARL:UT 76 Reuben H. Wallace, ARL:UT 77 Library, ARL:UT

Reserve, ARL:UT